

**DYNACOMP**

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**HARMONIC  
ANALYZER**



## HARMONIC ANALYZER

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1427 Monroe Ave.  
Rochester, NY 14618

### INTRODUCTION

HARMONIC ANALYZER is a frequency spectrum analysis package designed with the engineer in mind. It applies the concept of the Fast Fourier Transform (FFT) to an input data set to provide a frequency domain representation of the function approximated by that input data. HARMONIC ANALYZER is based on DYNACOMP's popular FOURIER ANALYZER, but includes special data handling features.

The user may save and recall both the input data file and previously calculated results. Data files may be loaded (from cassette, diskette, or disk), added to, deleted from, and generally edited. As with FOURIER ANALYZER, the input data may be plotted before the calculations are started. The frequency domain results may also be plotted as well as saved (to be later recalled for new plots). The analyses are easily repeated without re-entering the data. This is handy when you wish to test the sensitivity of the results to a particular data point.

One of the unique features of HARMONIC ANALYZER is that the data need not be equally spaced in "time." A special spline interpolation is performed just prior to the FFT calculation to convert the the unequally spaced data into a new "data" file which is uniformly spaced as required by the FFT.

### RELATION TO OTHER DYNACOMP FREQUENCY TRANSFORM PROGRAMS

DYNACOMP's FOURIER ANALYZER and TRANSFER FUNCTION ANALYZER were designed to be used on limited duration signals. The resulting transforms, therefore, represent a continuous spectrum. HARMONIC ANALYZER is meant to handle another class of signals; those which are repetitive and which, therefore, have discontinuous spectra. If FOURIER ANALYZER or TFA are used on single period waveforms belonging to repetitive wave-trains, the results will be wrong (not the fault of the program). Similarly, if HARMONIC ANALYZER is applied to limited duration signals, the results are very misleading. The truncated sinewave example given later demonstrates this.

## THEORY

Repetitive signals may be decomposed into sinusoidal components as follows:

$$f(t) = \sum_{n=-\infty}^{\infty} a_n e^{i\omega_n t}$$

where

$$i = \sqrt{-1}$$

$$\omega_n = n\omega_0 \quad (\text{radians/second})$$

$$\omega_0 = 2\pi/T_0 \quad (\text{radians/second})$$

$$T_0 = \text{period of signal (seconds)}$$

The Fourier Transform is defined as:

$$F(\omega) = \int_{-\infty}^{\infty} f(t) e^{-i\omega t} dt$$

Inserting the series representing  $f(t)$ , we get:

$$F(\omega) = a_n \quad (\omega = \omega_n)$$

$$= 0 \quad (\text{otherwise})$$

The Fourier Transform of repetitive signals is, therefore, discontinuous; composed of spikes.

As a theoretical example, consider the following function:

$$f(t) = \frac{1}{2} \quad -1 \leq t \leq 1$$

$$f(t) = 0 \quad \text{elsewhere}$$

This is not a repetitive waveform.

The associated Fourier Transform is a "sinc" function:

$$F(w) = \frac{\sin(w)}{w}$$

Note that this is a continuous function.

If instead the function were:

$$F(t) = \frac{1}{2} \quad -1 \leq t \leq 1$$

repeating at  $t=0, t=nT_0$

where  $T_0 \geq 2$

The spectrum would be discrete:

$$F_n = F(W_n) = \frac{\sin(W_n)}{W_n}$$

The larger  $T_0$  is, the closer together will be the  $W_n$ , and in the limit as  $T_0 \rightarrow \infty$  the spectrum will become continuous.

A classical example of the difference between the Fourier integral for limited duration signals and the DFT/FFT is the spectrum derived for a limited duration sinewave. The DFT and FFT algorithms generally give a spectrum with a simple spike at the sinewave frequency. This is what one would expect for a continuous (infinite in extent) sinewave. The Fourier integral instead presents a spectrum in which the signal energy is concentrated about the sinewave frequency, but with some spread (actually an offset "sinc" function).

## APPROXIMATION ERROR

By using only a finite number of data points to represent a (piece-wise) continuous function, we are introducing approximation error. In effect, the function is being digitized\*, or sampled. Thus, it is not possible to accurately and unambiguously capture the high frequency components of the original function. This limitation is described by the Nyquist sampling theorem which, in essence, states that the sampling rate (number of samples/unit time) should be at least twice the highest frequency component contained in the signal. For sampling rates below this, "aliasing" can occur. Spectrum calculations beyond this point should be considered suspect, though not necessarily invalid. Usually one avoids this problem by "filtering" the data before sampling. This may be done using the DATA SMOOTHER program also available through DYNACOMP, or DIGITAL FILTER (available February 1981). If the data input to HARMONIC ANALYZER is equally spaced, the sampling rate will be twice the highest frequency calculated, though perhaps not at or greater than twice the rate of the highest component in the original signal.

## THE NUMBER OF DATA POINTS

The FFT algorithm employed in HARMONIC ANALYZER requires that the number of data points processed be equal to an even power of 2 (e.g., 2, 4, 8, 16, 32, etc.). You will not be permitted to enter the transform step unless that is the case. The data file structure used by HARMONIC ANALYZER is generally the same as that employed by TFA (Transfer Function Analyzer). Thus, it is possible to load a data set which violates this condition, but data will have to be added or deleted to bring the number of points to an even power of two.

## NON-UNIFORM, DISORDERED DATA

Before HARMONIC ANALYZER enters the transform stage, a new data set is constructed from the one supplied. First, the "old" data set is ordered in terms of ascending "X" (or "t"). Next, the range of the N data points is determined and a new set of N equally spaced "X" (or "t") values chosen. Using a cubic spline interpolation, the "Y" (or "f") values corresponding to these equally spaced,

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\* Actually, digitization is a further source of error.

ordered values are then determined. This artificial data set is subsequently sent to the FFT subroutine. If the original data set was ordered and equally spaced, the artificial data set is just an image of the original data collection.

A cubic spline interpolation scheme was chosen over a cubic LaGrange (or Newton) in order to avoid the "ringing" associated with the latter.

#### USING HARMONIC ANALYZER TO CALCULATE FREQUENCY RESPONSE

HARMONIC ANALYZER may be directly applied to analyze the transfer function of a linear system. The experimental component is simple. A repetitive string of delta functions (e.g., pulse train of "spikes") is used as the input to the system to be analyzed. If the spike is transmitted unaltered, then  $F(w_n) = a$  constant. The Fourier transform of the delta function is unity. Therefore, if the spike is transmitted with distortion, the transform (as calculated by HARMONIC ANALYZER) of this distorted waveform is proportional to the frequency response of the system being tested. An example is given in the attached listings.

A word of caution is in order. Be careful that the input pulse is not so large that non-linear distortion occurs, or too wide. Ideally, the width should be less than the distance between sampling points.

#### PLOTTING WIDTH

The aspect ratio of the plotting routine is controlled by the parameter U. By increasing the value of U, fewer line feeds are inserted between plotted points. U is defined at the very beginning of the program and may be adjusted by the user to suit the terminal device employed.

In conclusion, HARMONIC ANALYZER is a fairly general and very powerful frequency analysis tool. To get the best results from its application, remember that it is a numerical approximation whose accuracy and resolution improves with more precise and extensive data.

## HARMONIC ANALYZER EXAMPLES

The following set of examples was generated using the North Star version of HARMONIC ANALYZER. All other versions are very similar in their behavior.

\*\*\* HARMONIC ANALYZER \*\*\*

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DYNACOMP  
P.O. BOX 142  
WEBSTER, NEW YORK 14580

THIS PROGRAM TRANSFORMS A GIVEN  
REPETITIVE DATA SET INTO ITS  
FREQUENCY DOMAIN REPRESENTATION  
USING THE FAST FOURIER TRANSFORM.

THE USER INPUTS THE NUMBER  
OF DATA POINTS TO BE TREATED.  
THIS NUMBER MUST BE AN EVEN POWER  
OF TWO. FOR EXAMPLE,  $N=2^K$ .  
THE PROGRAM WILL THEN PLOT  
THE DATA AND ITS SPECTRUM IN  
TERMS OF AMPLITUDE AND PHASE.

YOU MAY ALSO CHOOSE TO RECALL  
PREVIOUS DATA OR RESULTS.

DO YOU WISH TO PLOT  
PREVIOUS RESULTS (Y/N)? Y

INPUT NUMBER OF TABLE VALUES: 716

IS THE DATA TO BE LOADED  
FROM DISK (Y/N)? Y

INPUT THE DATA. YOU MAY EDIT IT LATER.

X( 1) = 71 Y( 1) = 70

X( 2) = 72 Y( 2) = 70

X( 3) = 73 Y( 3) = 70

X( 4) = 74 Y( 4) = 70

X( 5) = 75 Y( 5) = 70

X( 6) = 76 Y( 6) = 71

X( 7) = 77 Y( 7) = 71

X( 8) = 78 Y( 8) = 71

X( 9) = 79 Y( 9) = 71

X(10) = 710 Y(10) = 71

X(11) = 711 Y(11) = 71

X(12) = 714 Y(12) = 70

X(13) = 715 Y(13) = 70

X(14) = 716 Y(14) = 70

X(15) = 717 Y(15) = 70

X(16) = 712 Y(16) = 70

E.G., 2, 4, 8, 16, 32

The phase is restricted to the range 0-360°

A "Y" answer will lead to prompts concerning  
loading a previously saved "results" file.  
Control will then go directly to the plotting  
routines for amplitude and phase.

You will not be able to pass this point with  
an illegal number of data pairs.

If the data were to be loaded from disk (or  
cassette), the number of table values chosen  
above must be larger than the number of pairs  
in the file to be loaded. In this case we will  
build a 16 point data table and save it later.

At this stage data is entered until 16 pairs are  
recorded. If a mistake is made it can be corrected  
later. Note that the data need not be entered in  
"time" order. It will be sorted later.

The example we will first examine is a simple  
train of broad pulses.

This is an intentional error to be corrected.

Observe that this data point is out of order.







## CURRENT DATA FILE:

I	X(I)	Y(I)
1	1	0
2	2	0
3	3	0
4	4	0
5	5	0
6	6	1
7	7	1
8	8	1
9	9	1
10	10	1
CONTINUE		
11	11	1
12	12	0
13	13	0
14	14	0
15	15	0
16	16	0

We are going to repeat the analysis, but make a change in the input data set to test the interpolation procedure.

DO YOU WISH TO ADD TO DATA (Y/N): Y

SORRY, ARRAY SPACE FILLED.

DO YOU WISH TO DELETE DATA (Y/N): Y

SUGGESTION - DELETE STARTING AT THE END OF THE FILE.

WHICH DATA PAIR: 16

DO YOU WISH TO DELETE DATA (Y/N): Y

ERROR! THE CURRENT NUMBER OF DATA PAIRS IS NOT AN EVEN POWER OF 2.

DO YOU WISH TO ADD TO DATA (Y/N): Y

YOU MAY ADD 1 POINTS.

HOW MANY POINTS DO YOU WISH TO ADD: 1

X( 16 ) = 7.8 Y( 16 ) = 0

We can not add more data because the array has been filled. However, if we had answered the original "number of table values" question with 32, and had then load a data file from disk which contained only 16 points, more data could be added at this point.

When a data pair is deleted, the pairs above shift down one to fill the gap. Thus the "current data file" table shown above becomes incorrect above the pair deleted. If data pair 7 is deleted, pair 8 becomes pair 7, pair 9 becomes pair 8, and so on. In this case we deleted the last pair in the table as an example, and we will later replace it.

The program just checked the number of data pairs and found only 15. That is an invalid number for entry into the FFT. You will now be given a chance to correct that.

15 points currently exist. The array size initially chosen was 16. In this case you did not have much choice.

We put the 16<sup>th</sup> pair back.

DO YOU WISH TO DELETE DATA (Y/N): Y

DO YOU WISH TO EDIT ANY DATA PAIR (Y/N): Y

WHICH PAIR: 7

X( 7 ) = 7.8 Y( 7 ) = 1

Originally pair 7 was (7,1). We will shift that data point to (7.8,1). Note that it is now right next to point 8 which is at (8,1).

DO YOU WISH TO EDIT ANY DATA PAIR (Y/N): Y

SORTING DATA...

DO YOU WISH TO SAVE THE CURRENT DATA ON DISK (Y/N): Y

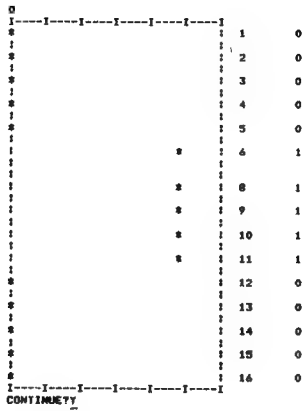
DO YOU WISH TO REMOVE THE DC COMPONENT (Y/N): Y

The last time we removed the DC component

PLOT DATA (Y/N): Y

WHAT IS THE PLOT WIDTH: 30

DISPLAY PLOT VALUES (Y/N): Y



Point 7 has moved so close to point 8 that it is coincident on the plot. Thus only point 8 is displayed. Note that we are still describing the same square pulse, but this time the points are not evenly spaced.

#### FOURIER TRANSFORM UNDER WAY....

FREQUENCY CYCLES/UNIT	AMPLITUDE SCALED	PHASE DEGREES
0	1	0
.0425	.7893	191
.125	.308	22
.1875	.1148	34
.25	.2357	225
.3125	.0767	56
.375	.1276	67
.4375	.157	259
.5	0	0

Since we included the DC term this time, we can not make an immediate comparison between the amplitude values calculated for the evenly spaced data and the last set. The amplitude results are always scaled such that the largest value is normalized. However, observe that the phase values are identical.

On to another example.

RUN

In this example we will see how a data file may be loaded from disk (or cassette) and more data added to it. This particular example is interesting in that the data is not artificial, but rather the result of an experiment. In this experiment a filter was probed with a train of pulses and the output displayed on an oscilloscope. The distorted output represents the "impulse response", and its transform the transfer function.

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THE DATA AND ITS SPECTRUM IN  
TERMS OF AMPLITUDE AND PHASE.

YOU MAY ALSO CHOOSE TO RECALL  
PREVIOUS DATA OR RESULTS.

DO YOU WISH TO PLOT  
PREVIOUS RESULTS (Y/N)? Y

INPUT NUMBER OF TABLE VALUES: 768

IS THE DATA TO BE LOADED  
FROM DISK (Y/N)? Y

WHAT IS THE FILE NAME: TRDAR\_TA

IS THE DISK READY? Y

40 POINTS WERE FOUND.

CAUTION! THE FILE CONTAINED  
A NUMBER OF DATA PAIRS WHICH IS  
NOT AN EVEN POWER OF 2.

The actual set of data consists of 40 data pairs.  
The next number higher than 40 which will be accepted  
is 64.

Typo's can usually be corrected before the "RETURN"  
or "ENTER" key is pressed by using "delete", "rubout",  
"back-arrow", or whatever the corresponding key is  
on your keyboard.  
Only 40 points were found.

The computer quickly becomes aware that the number  
of current data pairs is not acceptable.

CURRENT DATA FILE:

I	X(I)	Y(I)
1	.00002	0
2	.00004	.6
3	.00006	1.5
4	.00008	2.4
5	.0001	3
6	.00012	3.4
7	.00014	3.6
8	.00016	3.4
9	.00018	2.8
10	.0002	2.2
CONTINUE? <u>Y</u>		
11	.00022	1.5
12	.00024	.6
13	.00026	0
14	.00028	-.6
15	.0003	-1.2
16	.00032	-1.6
17	.00034	-1.8
18	.00036	-1.9
19	.00038	-2
20	.0004	-1.9
CONTINUE? <u>Y</u>		

```

21 .00042 -1.9
22 .00044 -1.8
23 .00046 -1.6
24 .00048 -1.5
25 .0005 -1.4
26 .00052 -1.3
27 .00054 -1.2
28 .00056 -1.1
29 .00058 -1
30 .0006 -.9
CONTINUE?Y

```

```

31 .00062 -.85
32 .00064 -.8
33 .001 -.1
34 .0013 .3
35 .002 .4
36 .0025 .35
37 .003 .25
38 .0035 .2
39 .004 .15
40 .0045 .1
CONTINUE?Y

```

Observe that for long time lengths and low amplitudes, the sampling interval has been increased.

DO YOU WISH TO ADD TO DATA (Y/N): YY

Two options exist. We can add data to get to 64 pairs, or we can delete data to get to the 32 level. The first option is chosen.

YOU MAY ADD 24 POINTS.

HOW MANY POINTS DO YOU WISH TO ADD: 724

X( 41) = 7.00003  
Y( 41) = 7.3

X( 42) = 7.00005  
Y( 42) = 71.06.5

X( 43) = 7.00007  
Y( 43) = 71.95

X( 44) = 7.00009  
Y( 44) = 72.7

24 "new" data points are added. They are just the averages of the first 25 points; we are inserting linearly interpolated data.

This part of the print-out is boring.

```

|           |
|           |
|           |

```

DO YOU WISH TO DELETE DATA (Y/N): YN

DO YOU WISH TO EDIT ANY DATA PAIR (Y/N): YN

SORTING DATA...

The sort takes some time for large data sets. remember that the sort is followed by interpolation to create an equally spaced "data" table.

DO YOU WISH TO SAVE THE CURRENT  
DATA ON DISK (Y/N): YY

WHAT IS THE FILE NAME: TRDATA

Recall that the file name is always required, but not actually used for the cassette versions.

IS THE DISK READY?Y

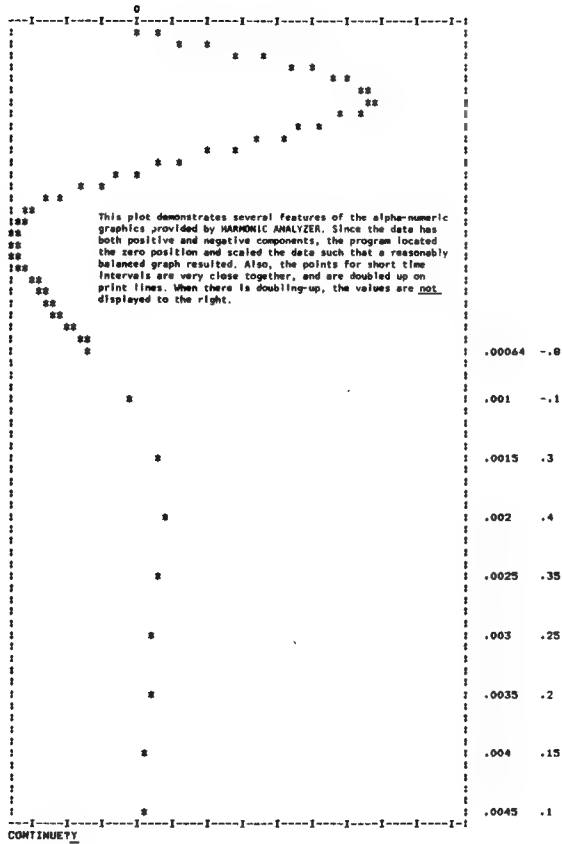
DO YOU WISH TO REMOVE THE  
DC COMPONENT (Y/N): YN

PLOT DATA (Y/N): YY

WHAT IS THE PLOT WIDTH: 765

Always try to use a plot width greater than the number of data pairs.

DISPLAY PLOT VALUES (Y/N): YY



## FOURIER TRANSFORM UNDER WAY....

FREQUENCY CYCLES/UNIT	AMPLITUDE SCALED	PHASE DEGREES
-----	-----	-----
0	.5604	0
219.7268	.9953	123
439.4535	1	44
659.1802	.9022	10
878.907	.8696	348
1098.6338	.849	329
1318.3606	.8086	312
1538.0873	.7527	297
1757.8141	.754	292
1977.5408	.8377	272
CONTINUE?Y		
2197.2676	.7869	248
2416.9944	.578	232
2636.721	.4942	225
2856.4478	.4439	218
3076.1746	.4023	209
3295.9014	.373	200
3515.628	.3398	188
3735.3548	.2605	168
3955.0816	.1748	181
4174.8082	.2069	188
CONTINUE?Y		
4394.535	.2198	163
4614.2618	.1533	152
4833.9885	.1081	152
5053.7153	.0896	142
5273.4421	.0907	172
5493.1688	.1055	170
5712.8955	.1212	145
5932.6223	.0419	113
6152.3491	.0409	222
6372.0757	.1	182
CONTINUE?Y		
6591.8025	.0726	161
6811.5293	.0506	161
7031.2559	.0431	180

DO YOU WISH AN AMPLITUDE  
VERSUS FREQUENCY PLOT (Y/N)? Y

WHAT IS THE PLOT WIDTH? 740

DISPLAY PLOT VALUES (Y/N)? Y

There are 33 transform pairs; a plot width of 40  
is sufficient.



	0	.56
#####	1	
#####	219.73	1
!		
#####	439.45	1
!		
#####	859.18	.9
!		
#####	878.91	.87
!		
#####	1098.63	.85
!		
#####	1318.36	.81
!		
#####	1538.09	.75
!		
#####	1757.81	.75
!		
#####	1977.54	.84
!		
#####	2197.27	.79
!		
#####	2416.99	.58
!		
#####	2636.72	.5
!		
#####	2856.45	.44
!		
#####	3076.17	.4
!		
#####	3295.9	.37
!		
#####	3515.62	.34
!		
#####	3735.35	.26
!		
#####	3955.08	.17
!		
#####	4174.8	.21
!		
#####	4394.53	.22
!		
#####	4614.26	.15
!!		
###	4833.98	.11
!		
##	5053.71	.09
!		
##	5273.44	.09
!!!		
!	5493.16	.11
!		
###	5712.89	.12
!		
!	5932.62	.04
!		
!	6152.34	.04
!		
####	6372.07	.1
!		
##	6591.8	.07
!		
!	6811.52	.05
!		
!	7031.25	.04
-----		

This filter actually had no DC response. Most likely there was a small offset in the scope trace from which the data was obtained. Knowing that there was no DC response we could have chosen to arbitrarily remove the DC component before the FFT calculation. That option was not chosen. If it had been, the amplitude would have been zero at zero hertz.

DO YOU WISH A PHASE PLOT (Y/N): TY

WHAT IS THE PLOT WIDTH: 740

DISPLAY PLOT VALUES (Y/N): TY

[illegible]



